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APPLICATION NO. FILING DATE FIRST NAMED INVENTOR ATTORNEY DOCKET NO. CONFIRMATION NO. 09/658,771 09/11/2000 Madhu Yarlagadda 17887-006000 4424 EXAMINER 20350 01/29/2004 TOWNSEND AND TOWNSEND AND CREW, LLP MEW, KEVIN D TWO EMBARCADERO CENTER ART UNIT PAPER NUMBER **EIGHTH FLOOR** SAN FRANCISCO, CA 94111-3834 2664 **DATE MAILED: 01/29/2004**

Response Due 4/29/04 538

Please find below and/or attached an Office communication concerning this application or proceeding.

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·		Application No.	Applicant(s)
Office Action Summary		09/658,771	YARLAGADDA ET AL.
		Examiner	Art Unit
		Kevin Mew	2664
The MAILING DATE of this communication appears on the cover sheet with the correspondence address Period for Reply			
A SHORTENED STATUTORY PERIOD FOR REPLY IS SET TO EXPIRE 3 MONTH(S) FROM THE MAILING DATE OF THIS COMMUNICATION. - Extensions of time may be available under the provisions of 37 CFR 1.136(a). In no event, however, may a reply be timely filed after SIX (6) MONTHS from the mailing date of this communication. - If the period for reply specified above is less than thirty (30) days, a reply within the statutory minimum of thirty (30) days will be considered timely. - If NO period for reply is specified above, the maximum statutory period will apply and will expire SIX (6) MONTHS from the mailing date of this communication. - Failure to reply within the set or extended period for reply will, by statute, cause the application to become ABANDONED (35 U.S.C. § 133). - Any reply received by the Office later than three months after the mailing date of this communication, even if timely filed, may reduce any earned patent term adjustment. See 37 CFR 1.704(b). Status			
1)⊠ Res _i	ponsive to communication(s) filed on 11.	September 2000	
2a) <mark>⊟ Th</mark> is	action is FINAL . 2b)⊠ Th	nis action is non-final.	
3) Since this application is in condition for allowance except for formal matters, prosecution as to the merits is closed in accordance with the practice under Ex parte Quayle, 1935 C.D. 11, 453 O.G. 213.			
Disposition of Claims			
4) Claim(s) 1-9 is/are pending in the application.			
4a) Of the above claim(s) is/are withdrawn from consideration.			
5) Claim(s) is/are allowed.			
6)⊠ Claim(s) <u>1-6 and 9</u> is/are rejected.			
7) Claim(s) <u>7, 8</u> is/are objected to.			
8) Claim(s) are subject to restriction and/or election requirement. Application Papers			
9)☐ The specification is objected to by the Examiner.			
10)⊠ The drawing(s) filed on <u>11 September 2000</u> is/are: a)⊠ accepted or b)□ objected to by the Examiner.			
Applicant may not request that any objection to the drawing(s) be held in abeyance. See 37 CFR 1.85(a).			
11)☐ The proposed drawing correction filed on is: a)☐ approved b)☐ disapproved by the Examiner.			
If approved, corrected drawings are required in reply to this Office action.			
12)☐ The oath or declaration is objected to by the Examiner.			
Priority under 35 U.S.C. §§ 119 and 120			
13) Acknowledgment is made of a claim for foreign priority under 35 U.S.C. § 119(a)-(d) or (f).			
a)☐ All b)☐ Some * c)☐ None of:			
1. Certified copies of the priority documents have been received.			
2. Certified copies of the priority documents have been received in Application No			
 3. Copies of the certified copies of the priority documents have been received in this National Stage application from the International Bureau (PCT Rule 17.2(a)). * See the attached detailed Office action for a list of the certified copies not received. 			
14) Acknowledgment is made of a claim for domestic priority under 35 U.S.C. § 119(e) (to a provisional application).			
 a) ☐ The translation of the foreign language provisional application has been received. 15)☐ Acknowledgment is made of a claim for domestic priority under 35 U.S.C. §§ 120 and/or 121. 			
Attachment(s)			
2) Notice of Dra	erences Cited (PTO-892) ftsperson's Patent Drawing Review (PTO-948) Disclosure Statement(s) (PTO-1449) Paper No(s) <u>5</u>	5) Notice of Informa	ary (PTO-413) Paper No(s) al Patent Application (PTO-152)

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Claim Rejections - 35 USC § 102

1. The following is a quotation of the appropriate paragraphs of 35 U.S.C. 102 that form the basis for the rejections under this section made in this Office action:

A person shall be entitled to a patent unless -

(e) the invention was described in (1) an application for patent, published under section 122(b), by another filed in the United States before the invention by the applicant for patent or (2) a patent granted on an application for patent by another filed in the United States before the invention by the applicant for patent, except that an international application filed under the treaty defined in section 351(a) shall have the effects for purposes of this subsection of an application filed in the United States only if the international application designated the United States and was published under Article 21(2) of such treaty in the English language.

Claim 9 is rejected under 35 U.S.C. 102(e) as being anticipated by Ladd et al. (US Patent 6,269,336).

Ladd discloses a communication node (see element 212, Figure 3) that comprises of Voice Recognition (VRU) Client, Voice Recognition (VRU) server, and VoIP unit for processing requests from user interactively based upon the information services desired by the user and returns the corresponding information to the user (integrated VoIP network comprising a plurality of voice processing modules for processing requests without forwarding voice data to an end destination, see lines 4-67, col. 4, lines 1-19, col. 5 and elements 212, 232, 234, 252, 254, 256, 248, Figure 3).

Claim Rejections - 35 USC § 103

The following is a quotation of 35 U.S.C. 103(a) which forms the basis for all obviousness rejections set forth in this Office action:

(a) A patent may not be obtained though the invention is not identically disclosed or described as set forth in section 102 of this title, if the differences between the subject matter sought to be patented and the prior art are such that the subject matter as a whole would have been obvious at the time the invention was made to a person having ordinary skill in the art to which said subject matter pertains. Patentability shall not be negatived by the manner in which the invention was made.

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2. Claims 1-6 are rejected under 35 U.S.C. 103(a) as being unpatentable over Ladd et al. in view of Neubauer (US Patent 6,269,095).

Regarding claims 1 & 6, Ladd discloses a method in which the VoIP unit in a telecommunication node (telephone application services using a managed VOIP network, see lines 55-58, col. 9; note that telecommunication node 212 is the managed VoIP network) allows users to access information via the internet using voice inputs and commands, where speech inputs received at the VoIP unit and is converted to a VoIP protocol that would be transmitted over the internet (voice data transmitted over the network is codified in a native VOIP format, see lines 55-66, col. 9). The said method disclosed by Ladd comprises of:

a telephone switch (see element 230, Figure 3), preferably connected to the Voice Recognition (VRU) client and the VoIP unit, receives incoming calls from the communication device (providing a plurality of channels for handling incoming telephone calls, see element 204, Figure 3) routed over the internet via the VoIP unit; and a VRU client, including a random access memory (shared memory, see line 17, col. 8), is capable of connecting individual telephone lines directly to the switch (shared memory accessible to all channels, see lines 13-17, col. 8 and element 230, Figure 3);

receiving a user call into the electronic network (receiving a first incoming telephone call service requestor over the managed VOIP network, see lines 6-7, col. 4 and element 104, Figure 3) that can be routed over the internet via the VoIP unit. As it is well-known in the art teaching that one the underlying protocol of the internet is IP, and the fact that Ladd discloses the VoIP unit is capable of receiving and converting speech inputs (voice data) to a VoIP protocol

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that can be transmitted over the internet (see lines 55-66, col. 9), it is inherent the VoIP protocol would be encapsulated in the IP packets received;

establishing a connection or a link to the destination of the information source desired by the user (setting up a connection between the incoming telephone call and a first one of said channels for handling the incoming telephone call, see lines 40-43, col. 4);

processing the information retrieved from the destination of the information source and providing an output such as a speech communication response to the user based upon the retrieved information (identifying request service, see lines 50-53, col. 4, and accessing response voice data responsive to the requested service, see lines 50-56, col. 4);

As it is well-known in the art teaching that the underlying protocol of the internet is IP, and the fact that Ladd discloses the VoIP unit is capable of receiving and converting speech inputs (voice data) to a VoIP protocol that can be transmitted over the internet (see lines 55-66, col. 9), it is inherent the response speech communication (response voice data) would be encapsulated in the IP packets when transmitted over said managed VoIP network to the user (encapsulating said response voice data in a second plurality of response IP packets; and sending said plurality of response IP packets over said managed VoIP network to the service requestor).

Regarding claim 6, Ladd further discloses reading and decoding caller line identification (CLI) of the call and determine whether the CLI of the call is found in a stored CLI list of subscribers (extracting calling ID line data from VOIP call signaling protocol to obtain location information about the service requestor, see lines 6-15, col. 4). In addition, Ladd

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discloses providing an announcement to the caller (accessing customized voice data, see lines 19-21, col. 4) after the caller is identified.

Regarding claims 1 & 6, Ladd does not explicitly show the response voice data would be stored in the native VoIP format in a shared memory.

However, Neubauer discloses that it is well-known in the art that a VoIP gateway for voice transmission on the internet, configured to provide IP call control and IP data transport, provides compression/decompression of voice channels using G.723.1 vocoding (see lines 11-19, col. 1).

Therefore, it would have been obvious to one ordinary skill in the art at the time the invention was made to modify the VoIP unit of Ladd such that the such as the voice response data would be stored in native VoIP format such as the VoIP format G.723.1 taught by Neubauer. Having provided a memory means coupled to the VoIP unit for storing voice response data in G.723.1 format would modify VoIP unit of Ladd. The motivation to do so is to provide store voice response data in a compressed format because it would reduce the storage capacity for storing voice data and thereby increase I/O read/writes and the overall interactive voice response system performance.

Regarding claims 2 & 3, Ladd discloses that speech inputs (voice data) are received at the VoIP unit and are converted to a VoIP protocol that would be transmitted over the internet (voice data transmitted over the network is codified in a native VOIP format, see lines 55-66, col.

9). Also, it is well-known in the art teaching that the underlying protocol of the internet is IP that it is inherent response speech communication (response voice data) would be encapsulated in the IP packets when transmitted over said managed VoIP network to the user. By the same art

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teaching, it is inherent that the IP packets received would be processed in order to extract the response speech communication out of the IP packets (extracting voice data from said received IP packets and processing voice data in native format). Furthermore, Ladd discloses said communication node (see element 212, Figure 3) would receive speech commands or DTMF tones (identifying a DTMF signal and determining a rerquested service associated with an identified DTMF signal, see lines, 17-19, col. 5) from the user to retrieve desired information, and utilize speaker verification to identify a particular speech pattern of the user (performing speech analysis on extracted voice data to identify the service requested, lines 55-57, 60-62, col. 6).

Regarding claim 4, Ladd discloses a VRU server receives speech communications from the user and compares the speech communications against a vocabulary in the database server unit, and then provides a specific function associated with the output signal. If the function requires text to speech (TTS) conversion, the TTS unit would be called upon to process the textual data and converts the data to voice data (accessing voice response data and determining whether said requested service requires text to speech (TTS) conversion, if so invoking a TTS module that converts text to non-native voice data not in native VoIP format, see lines 55-67, col. 8, and lines 1-10, col. 9).

Ladd does not explicitly show that voice data to non-native voice data and then from nonative voice data to native VoIP format.

However, Neubauer discloses that it is well-known in the art that a VoIP gateway for voice transmission on the internet, configured to provide IP call control and IP data transport,

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provides compression/decompression of voice channels using G.723.1 vocoding (native VoIP format, see lines 11-19, col. 1).

Therefore, it would have been obvious to one ordinary skill in the art at the time the invention was made to modify the TTS unit of Ladd such that the voice response data being accessed would be stored in native VoIP format, such as the VoIP format G.723.1 taught by Neubauer, before the TTS unit is used to convert text into speech for transmission to the caller. Having provided a vocoder means coupled to the TTS unit for converting voice response data from non-native format to native VoIP G.723.1 format would modify TTS unit of Ladd. The motivation to do so is to convert voice response data in a compressed format because it would reduce the storage capacity for storing voice data and thereby increase I/O read/writes and the overall interactive voice response system performance.

Regarding claim 5, Ladd discloses a VRU server receives speech communications from the user and compares the speech communications against a vocabulary in the database server unit, and then provides a specific function associated with the output signal. If the function requires a speech recognition module, the automatic speech recognition (ASR) unit of the VRU server would be called upon to associate a specific function associated with the recognized voice pattern (accessing voice response data and determining whether received voice data will be processed by a speech recognition module, see lines 27-39, col. 9).

Ladd does not explicitly show that native VoIP format would be converted to non-native voice data prior to speech recognition by the ASR unit.

However, Neubauer discloses that it is well-known in the art that a VoIP gateway for voice transmission on the internet, configured to provide IP call control and IP data transport,

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provides compression/decompression of voice channels using G.723.1 vocoding (see lines 11-19, col. 1).

Therefore, it would have been obvious to one ordinary skill in the art at the time the invention was made to modify the ASR unit of Ladd such that the voice response data stored in native VoIP format, such as the VoIP format G.723.1 taught by Neubauer, would be converted to non-native format prior to speech recognition. Having provided a vocoder means coupled to the ASR unit for converting response voice data from native VoIP G.723.1 format into non-native format prior to speech recognition would modify the ASR unit of Ladd. The motivation to do so is to convert voice response data from a compressed native VoIP format to a decompressed non-native format because the voice data would have to be converted into a particular non-native form supported only by the corresponding ASR unit.

Allowable Subject Matter

3. Claims 7 & 8 are objected to as being dependent upon a rejected base claim, but would be allowable if rewritten in independent form including all of the limitations of the base claim and any intervening claims.

Regarding claim 7, the method of claim 1 further comprising the act of providing an I/O thread for each channel managing all I/O, with I/O thread performing the following acts:

while playing a message, giving higher priority to data transmission than to data reception, and while recording a message, giving higher priority to data reception than to data transmission.

Regarding claim 8, the method of claim 1 further comprising the acts of:

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providing a plurality of message access servers for controlling access to shared memory;

and

utilizing a service requestor ID to access a user database holding an association between the ID and a home MAS for accessing response data for the service requestor.

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Conclusion

4. The prior art made of record and not relied upon is considered pertinent to applicant's disclosure with respect to integrated VoIP system.

US Patent 6,584,110 to Mizuta et al.

US Patent 6,141341 to Jones et al.

US Patent 5,884,262 to Wise et al.

US Patent 6,456,967 to Yeom et al.

US Patent 6,446,127 to Schuster et al.

US Patent 6,654,456 to Mandalia et al.

5. Any inquiry concerning this communication or earlier communications from the examiner should be directed to Kevin Mew whose telephone number is 703-305-5300. The examiner can normally be reached on 9:00 am - 5:30 pm.

If attempts to reach the examiner by telephone are unsuccessful, the examiner's supervisor, Ricky Ngo can be reached on 703-305-4798. The fax phone number for the organization where this application or proceeding is assigned is (703) 872-9306.

Any inquiry of a general nature or relating to the status of this application or proceeding should be directed to the receptionist whose telephone number is 703-305-3900.

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